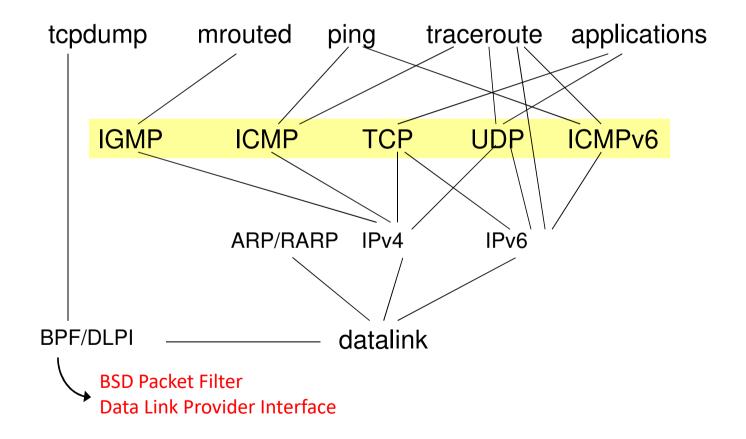
Network Programming: Ch.2 Transport Layer: TCP and UDP Li-Hsing Yen NYCU Ver. 1.0.1

### Transport Layer: TCP and UDP

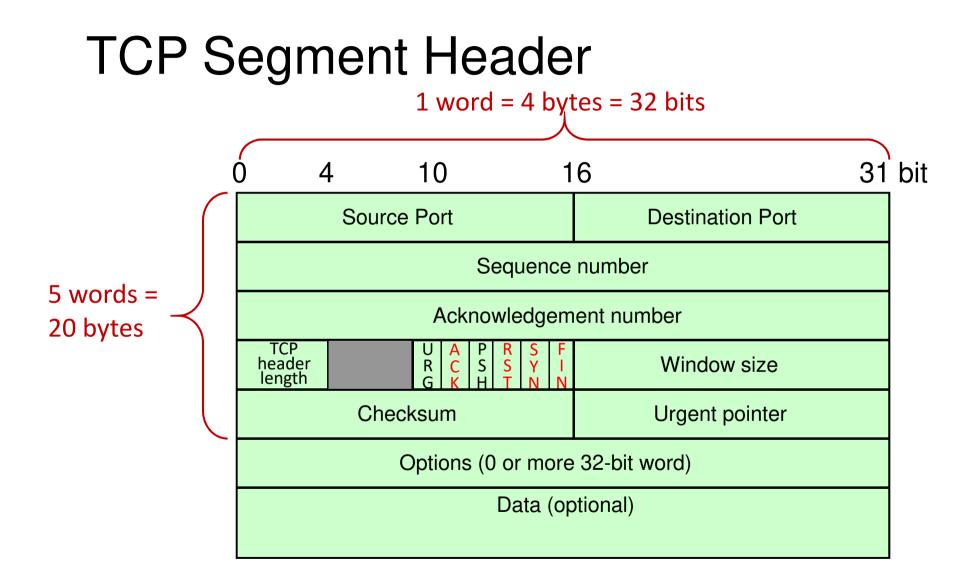
- Overview of TCP/IP protocols
- Comparing TCP and UDP
- TCP connection: establishment, data transfer, and termination
- Allocation of port numbers
- Size matters: MTU, datagram, MSS, buffer
- Standard Internet services and applications
- Debugging techniques and tools

#### **Overview of TCP/IP Protocols**



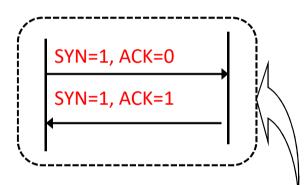
## Comparing TCP and UDP

	ТСР	UDP
Binding between client and server	Yes (connection- oriented)	No (connection-less)
Data	Byte-stream	Record
Reliability	Yes (ack, time- out, retx)	No
Sequencing	Yes	No
Flow control	Yes (window- based)	No
Full-duplex	Yes	Yes



## Some TCP Header Fields

- Sequence number (32 bits)
  - actually the byte number
- Acknowledgement number
  - specifies the next byte expected



• Fl	lags
------	------

URG	1 if the Urgent pointer is in use	RST	To reset a connection	
ACK	the Acknowledgement number is valid	SYN	To denote connection request and accepted	
PSH	This segment requests PUSH	FIN	To release a connection (in one direction)	

# More TCP Header Fields

#### • Window size (16 bits)

 tells how many bytes may be sent starting at the byte acknowledged (receiver's window size)

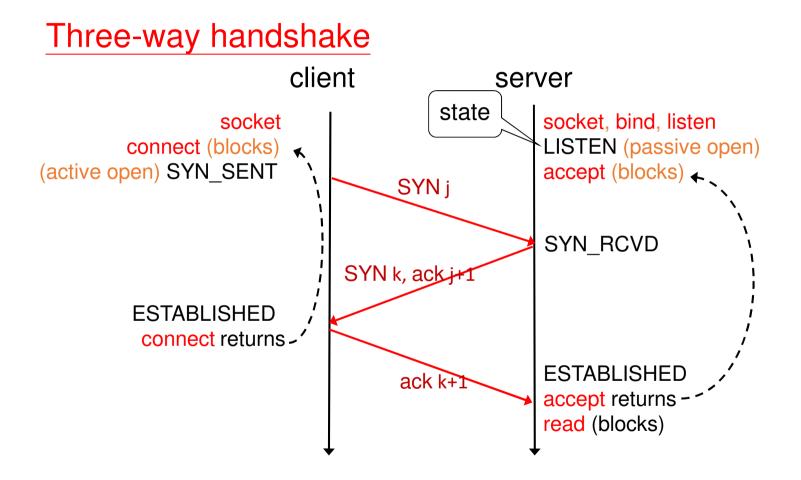
#### • Urgent pointer (16 bits)

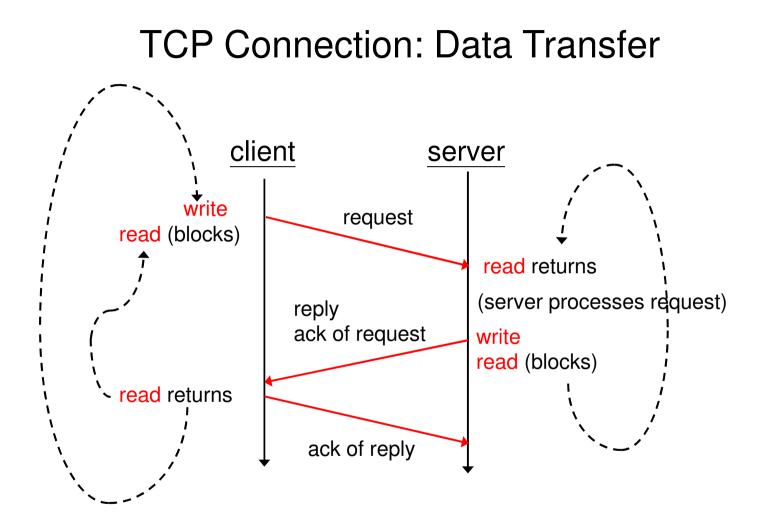
- indicate a byte offset from the current sequence number at which urgent data are to be found
- urgent data: when an interactive user hits the DEL or CTRL-C key, the sending application puts some control information in data stream

# **TCP** Options

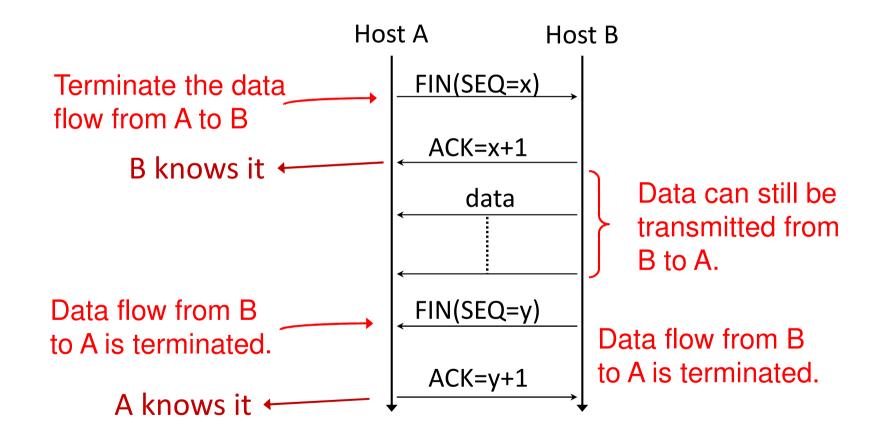
- MSS Option: maximum segment size
- Window scale: (new)
  - The maximum window size that either TCP can advertise to the other is 65535
  - This option shifts the advertised window by 0-14 bits (resulting in a maximum of 1GB)
- Timestamp: (new)
  - Prevent possible data corruption

#### **TCP** Connection: Establishment



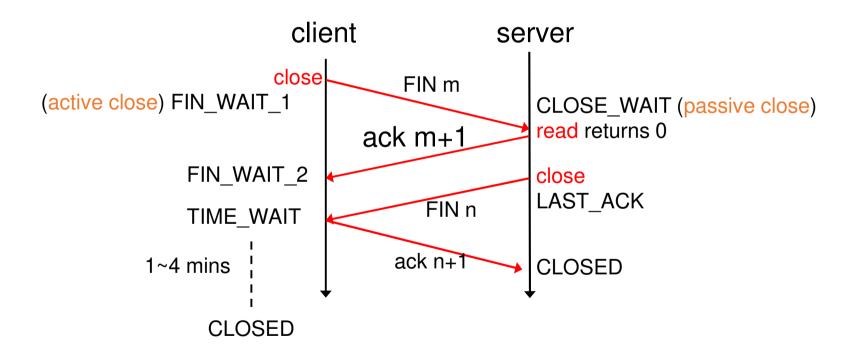


# **TCP Connection Release**

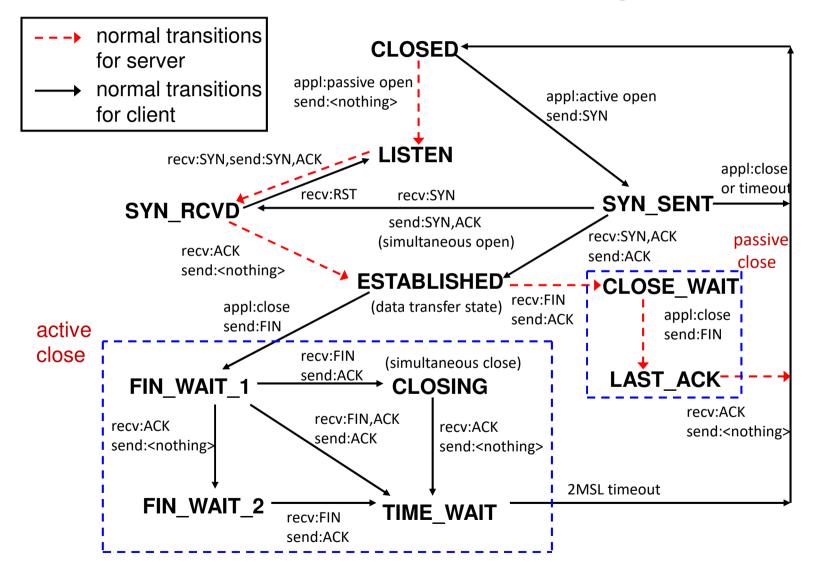


#### **TCP** Connection: Termination

#### Four-way handshake



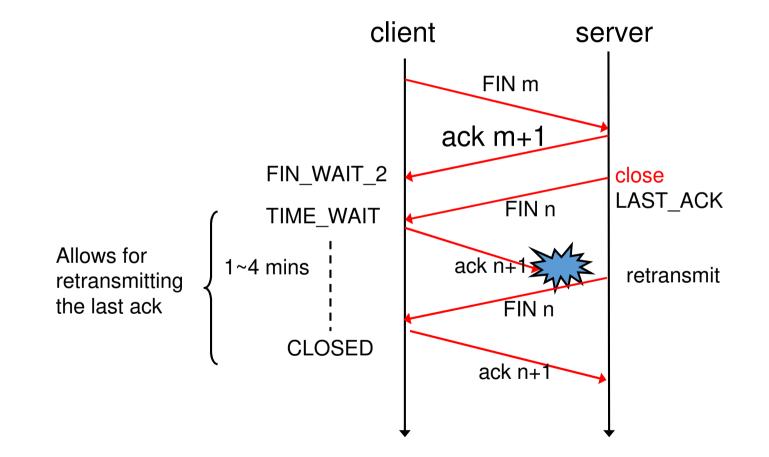
#### **TCP State Transition Diagram**



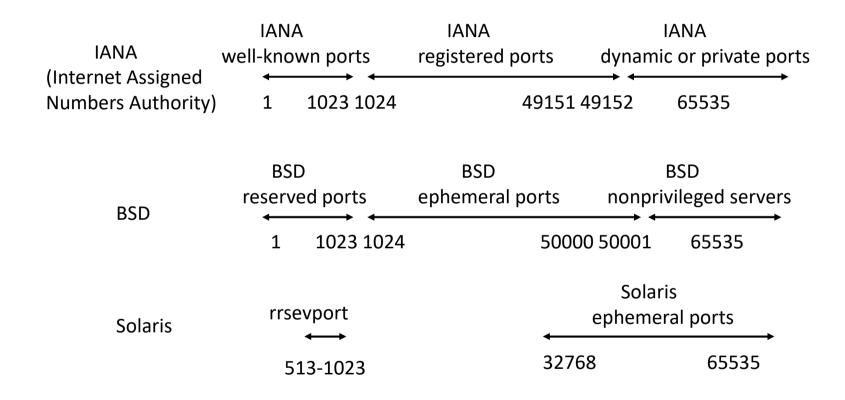
## TIME\_WAIT State

- Two reasons
  - To implement TCP's full-duplex connection termination reliably
    - The final ACK might need to be retransmitted
  - To allow old duplicate segments to expire in the network
    - Otherwise old duplicates will be misinterpreted as belonging to a new incarnation (having the same IP/port pair as the terminated one)

### Illustrating the First Reason



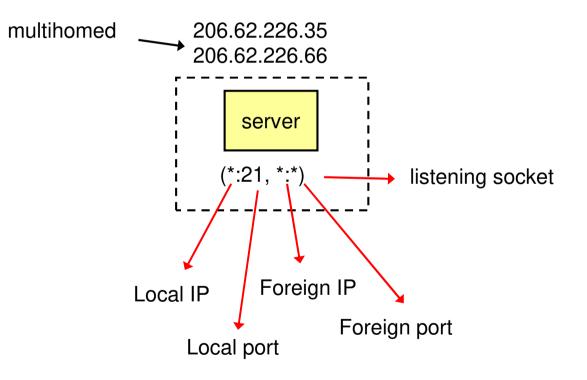
#### Allocation of Port Numbers



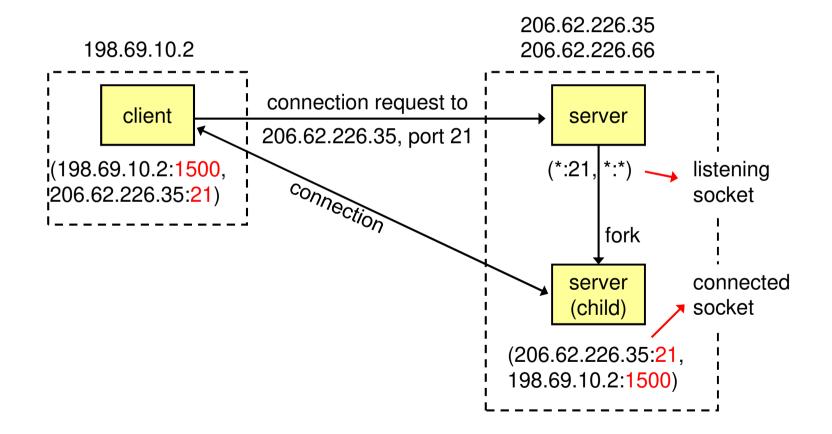
### Socket Pair

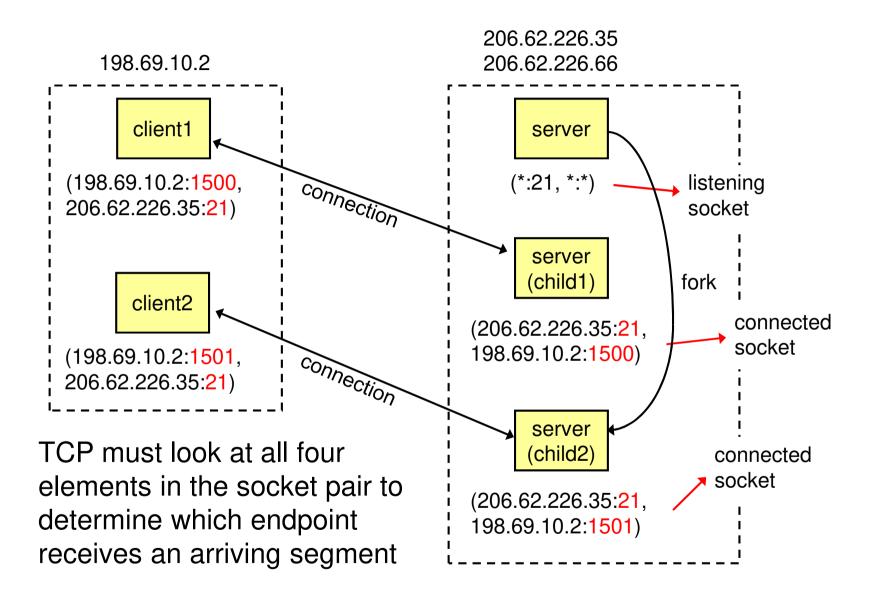
- 4-tuple that defines the two endpoints of a connection
  - The local IP address, local TCP port, foreign IP address, and foreign TCP port
- Uniquely identifies every TCP connection on the Internet

### A Concurrent Server



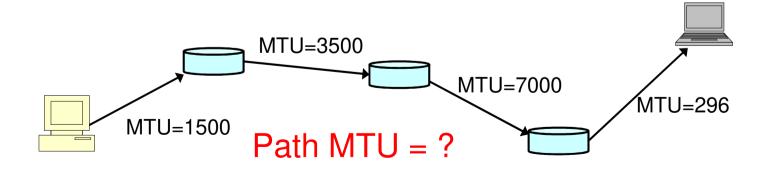
#### **Connect Request from Client to Sever**





#### **Buffer Sizes and Limitations**

- Link MTU (maximum transmission unit): Ethernet MTU: 1500 bytes, PPP MTU: configurable
- Path MTU: the smallest link MTU in the path, can be discovered by IPv4 DF (don't fragment) bit



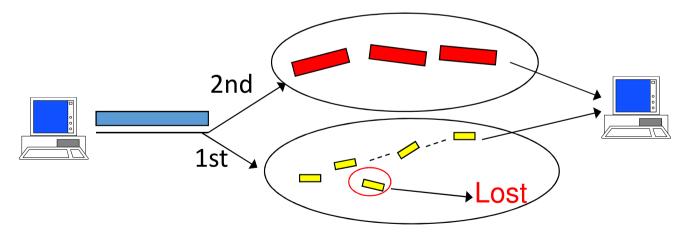
### Fragmentations

- Cut datagrams into fragments
- Performed when size of datagrams > link MTU
- Fragments are reassembled only at the final destination
- IPv4 DF bit tells routers not to fragment this datagram
  - May get ICMP "destination unreachable, fragmentation needed but DF bit set" error message

### **TCP** Fragmentation

• A segment will be fragmented when it passes through a transit network with a smaller MTU

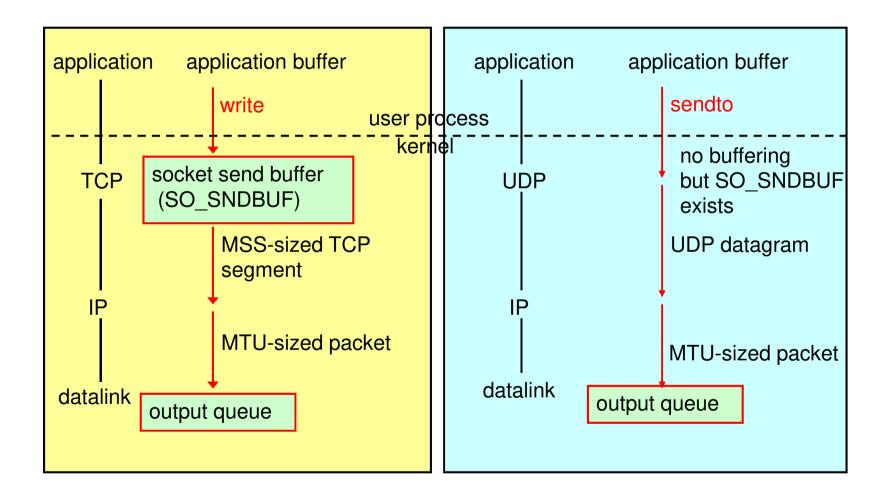
Problem with fragmentation



# Maximum and Minimum

- Maximum IP datagram: 65535 (IPv4), 65575 (IPv6) (IPv6 has 32-bit jumbo payload option)
- minimum IP reassembly buffer size: that we are guaranteed any implement must support
- TCP MSS (maximum segment size): actual value of reassembly buffer size, often the link MTU minus IP and TCP headers, to avoid fragmentation

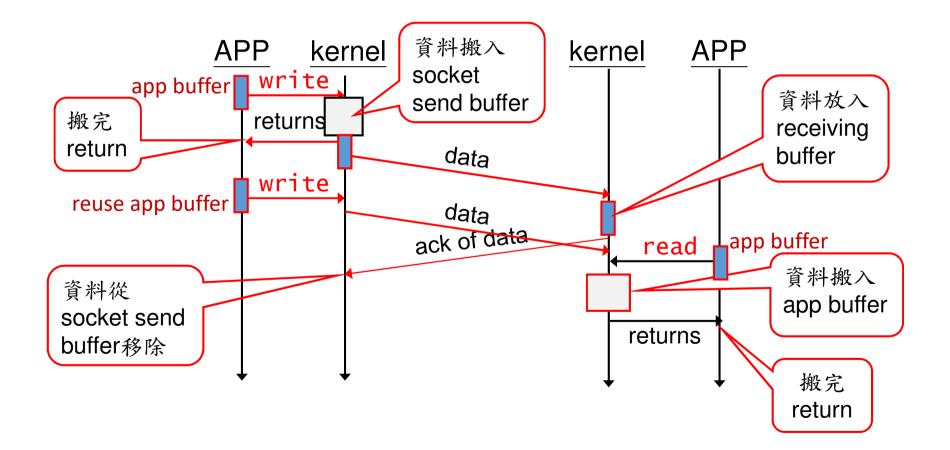
#### TCP Output and UDP Output



# TCP Output

- The write blocks if no room in the socket send buffer
- The kernel will not return from the write until the final byte in the application buffer has been copied into the socket send buffer
- Successful return from a write only tells us that we can reuse our application buffer
- TCP (kernel) keeps a copy of our data until it is acknowledged by the peer

### **TCP** Output Illustrated



# **UDP** Outputs

- UDP socket has a send buffer size but no real buffer
- If application's datagram > send buffer size, an error message is returned
- UDP need not keep a copy of our data
- A successful return from a write to a UDP socket tells us the datagram has been added to the data link output queue
- An error message will be generated if there is no room in the data link output queue



#### Standard Internet Services and Applications

- Standard services provided by *inetd* daemon: echo/port7/RFC862, discard/port9/RFC863, daytime/port13/RFC867, chargen/port19/RFC864, time/port37/RFC868
- tested by "telnet machine service", service mapped by /etc/services
- Common application types: diagnostic, routing protocol, datagram, virtual circuit, etc.

#### Protocol Usage of Various Common Applications

Application	IP	ICMP	UDP	ТСР
Ping		Х		
Traceroute		Х	Х	
OSPF	Х			
RIP			Х	
BGP				х
BOOTP			Х	
DHCP			Х	
NTP			Х	
TFTP			Х	
SNMP			Х	
SMTP				Х
Telnet				х
FTP				х
НТТР				х
NNTP				х
DNS			Х	Х
NFS			Х	х
RPC			Х	x

#### Debugging Techniques and Tools

- System call tracing: *truss* (in SVR4), *ktrace* & *kdump* (in BSD) (Note that socket is a system call in BSD, while putmsg and getmsg are the actual system calls in SVR4)
- sock developed by W.R. Stevens: used to generate special case conditions, as stdin/stdout client, stdin/stdout server, source client, sink server
- *tcpdump*: dump packets matching some criteria
- *netstat*: status of interfaces, multicast groups, per-protocol statistics, routing table, etc.
- *Isof* (list open files): which process has a socket open on a specified IP address or port